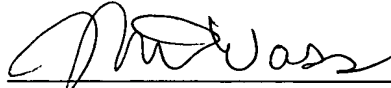


substitution of claims 15-39 are believed to put the present application in better form for review by the Examiner.

Respectfully submitted,

A handwritten signature in cursive script, appearing to read "M. Wasson", written over a horizontal line.

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The distribution function of speech is modeled as a GMM of time-correlated samples, leading to a distribution function for the speech spectral amplitude $s(f)$ as shown in Equation 2, where $\delta(s)$ is a one-sided Dirac delta function. The first term on the right hand side (RHS) of Equation 2 represents a signal of zero power, thus capturing the possibility that no signal of interest is present. The components of the summation in the second term on the RHS of Equation 2 are the components of the GMM model for the speech distribution function.

Equation 2

$$f_s(s) = (1 - q_s) \delta(s) + q_s \left\{ 2s \sum_i \frac{a_i}{\rho_i} \text{Exp}(-s^2 / \rho_i) \right\}$$

This speech model has two sets of frequency band dependent parameters which are dynamically updated during the processing, $\{P_s(f)\}$ and $\{q_s(f)\}$. The first is the a priori PSD of the speech, assuming that a speech signal is present at the frequency and time of interest. The second parameter is the a priori probability of a speech signal being present at that frequency and time. The speech distribution function also has a number of added parameters, $\{a_i\} = \{a_1, a_2, \dots, a_N\}$ and $\{\rho_i^\circ\} = \{\rho_1^\circ, \rho_2^\circ, \dots, \rho_N^\circ\}$. The $\{a_i\}$ are the weights of the N Gaussian components of the GMM, and the $\{\rho_i^\circ\}$ are the powers of each component when the speech PSD is normalized to $P_s(f) = 1$. In practice, $P_s(f)$ and $\{\rho_i^\circ\}$ are combined into a parameter set denoted as $\{\rho_i(f)\}$, where $\rho_i(f) = \rho_i^\circ P_s(f)$.

While both the $P_s(f)$ and $q_s(f)$ are dynamically updated during the processing, the $\{a_i\}$ and $\{\rho_i^o\}$ are determined from prior "training" to optimize processing results as averaged over a representative body of training data. This may typically be
 5 done by minimizing the mean-squared-error (MSE) between noise
free signals and the results from processing noisy input signals
based on those signals by mixing with varying types and levels
of interfering noise. The present invention may typically use
 five GMM components (denoted GMM5). However, more or less than
 10 five components can be employed. In addition, the $\{a_i\}$ may be
 further parameterized by the values of other key quantities,
 including but not limited to signal-to-noise ratio (SNR), which
 are adaptively and dynamically updated throughout the processing.
This may typically be done by determining different GMM model
 15 parameter values (the $\{a_i\}$ and $\{\rho_i^o\}$) versus SNR based on training
for different input SNRs, and interpolating between these model
parameter values based on the adaptively estimated input SNR
during the processing. One prior training of a GMM5 leads to a
 model for the speech distribution as shown in Figure 1 for $q_s =$
 20 0.5. Also shown is the corresponding distribution function for
 a Gaussian speech model with $q_s = 1$. For presentation purposes,
 the vertical axis is actually the distribution function for
 speech spectral power, which is simply $f(s^2/P_s)$, and the
 horizontal axis is $(s^2/P_s)^{1/2}$.

25 Noise PSD updating is mainly based on the following.
 Given a priori distribution functions for the noise and speech
 spectral amplitudes, and a new measurement of the noisy signal
 spectral amplitude, $r(f)$, a determination is made as to a best

The form of this estimator is depicted in Figures 4a and 4b. In these figures, the vertical axis is $(\langle s^2 | r \rangle / P_N)^{1/2}$, and the horizontal axis is $(r^2 / P_N)^{1/2}$. GMM5 results are given for 5 different SNRs, a nominal speech distribution function at $q_s = 0.5$, and as compared with a Gaussian speech model at $q_s = 1.0$, and also an extended Gaussian modes at $q_s = 0.5$. GMM5 results are in solid lines and Gaussian models are shown as dashed lines.

In a manner similar to the previous explanation, the speech spectral amplitude can also be estimated as follows.

Equation 13 ~~the equation has been changed~~

$$\langle \bar{s} | \bar{r} \rangle = \frac{q_s \sum_i \frac{a_i s_i}{(1+s_i)^2} \text{Exp}[(\frac{r^2}{P_N})(\frac{s_i}{1+s_i})]}{(1-q_s) + q_s \sum_i \frac{a_i}{1+s_i} \text{Exp}[(\frac{r^2}{P_N})(\frac{s_i}{1+s_i})]} \bar{r}$$

Change the
vector in
then 3 places

Note that in the special case with only one GMM component in the speech distribution function, and also with $q_s = 1$, the above expression reduces to a conventional Wiener filter.

For a typical set of GMM parameters, and at $q_s = 0.5$, and for different SNRs, the form of this estimator is shown in Figures 5a and 5b, where it is also compared with a Wiener filter at $q_s = 1.0$, and also with an extended Wiener filter based on a Gaussian speech model but with $q_s = 0.5$. In the figures, the vertical axis is $\langle s | r \rangle / (P_N)^{1/2}$, and the horizontal axis is $(r^2 / P_N)^{1/2}$.

It is further noted that the availability of separate estimates for both the speech spectral amplitude $\langle s | r \rangle$ and the